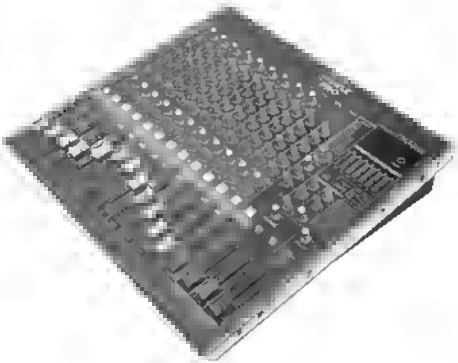
PYLE®
PRO



PMX-1609

16 Channel Stereo Console Mixer

OWNER'S MANUAL

Precautions

—For safe operation—

⚠WARNING

Installation

- Connect this unit's AC power adaptor only to an AC outlet of the type stated in this Owner's Manual or as marked on the unit. Failure to do so at a fire and electrical shock hazard.
- Do not allow water to enter this unit or allow the unit to become wet. Fire or electrical shock may result.
- Do not place a vase or any such liquid or small metal objects on top of this unit. Liquid or metal objects inside this unit are a fire and electrical shock hazard.
- Do not place heavy objects, including this unit, on top of the power cord. A damaged power cord is a fire and electrical shock hazard. In particular, be careful not to place heavy objects on a power cord covered by a carpet.

Operation

- Do not scratch, bend, twist, pull, or heat the power cord. A damaged power cord is a fire and electrical shock hazard.
- Do not remove the unit's cover. You could receive an electrical shock. If you think internal inspection, maintenance, or repairs is necessary, contact your dealer.
- Do not modify the unit. Doing so is a fire and electrical shock hazard.
- If lightning begins to occur, turn off the power switch of the unit as soon as possible, and unplug the power plug from the electrical outlet.
- If there is a possibility of lightning, do not touch the power plug while it is still connected. Doing so may be an electrical shock hazard.
- Use only the included AC power adaptor (PA-30) for this unit. Using other types may be a fire and electrical shock hazard.

In case an abnormality occurs during operation

- If the power cord is damaged (i.e., cut or bare wire is exposed), ask your dealer for a replacement. Using the unit with a damaged power cord is a fire and electrical shock hazard.
- Should this unit and AC adaptor be dropped or the cabinet be damaged, turn the power switch off, remove the power plug from the AC outlet, and contact your dealer. If you continue using the unit without heeding this instruction, fire or electrical shock may result.
- If you notice any abnormality, such as smoke, odor, or noise, or if a foreign object or liquid gets inside the unit, turn it off immediately. Remove the power plug from the AC outlet. Consult your dealer for repair. Using the unit in this condition is a fire and electrical shock hazard.

⚠CAUTION

Installation

- Keep this unit away from the following locations:
  - Locations exposed to oil splashes or steam, such as near cooking stoves, humidifiers, etc.
  - Unstable surfaces, such as a wobbly table or slope.
  - Locations exposed to excessive heat, such as inside a car with all the windows closed, or places that receive direct sunlight.
  - Locations subject to excessive humidity or dust accumulation.
- Hold the power plug when disconnecting it from an AC outlet. Never pull the cord. A damaged power cord is a potential fire and electrical shock hazard.
- Do not touch the power plug with wet hands. Doing so is a potential electrical shock hazard.
- To relocate the unit, turn the power switch off, remove the power plug from the AC outlet, and remove all connecting cables. Damaged cables may cause fire or electrical shock.

Operation

- Do not cover or wrap the AC power adaptor with a cloth or blanket. Heat may build up under the cloth or blanket, melting the case, or causing fire. Use only in a well-ventilated environment.
- If you know you will not use this unit for a long period of time, such as when going on vacation, remove the power plug from the AC outlet. Leaving it connected is a potential fire hazard.

—For correct operation —

Connector pin assignments

- XLR-type connectors are wired as follows:
  Pin 1: ground, Pin 2: hot (+), Pin 3: cold (–)
- INSERT TRS phone jacks are wired as follows:
  Sleeve: ground, Tip: send, Ring: return.

Replacement of Consumable Parts

- The performance of components with movable contacts—such as switches, rotary controls, faders, and connectors—deteriorates over time. While the rate of wear may vary greatly according to usage conditions, some amount of wear is unavoidable. When parts wear out, consult your dealer about appropriate replacements.

Interference from Cell Phones

- Use of a mobile phone near this unit may induce noise. If noise occurs, move the phone further from the unit.

- Always turn the power off when the mixer is not in use.
- Even when the power switch is in the "STANDBY" position, electricity is still flowing to the mixer at the minimum level. When you are not using the mixer for a long time, make sure you unplug the AC power adapter from the wall AC outlet.

Introduction

Thank you for your purchase of the PYLE PRO PMX1609 mixing console. This mixing console combines ease of operation with support for multiple usage environments, and is ideal for SR setups, installed systems, and many other such applications.

Please read through this Owner's Manual carefully before beginning use, so that you will be able to take full advantage of this mixer's superlative features and enjoy trouble-free operation for years to come.

Features

- The PMX1609 provides 16 input channels and mixes the signals into Stereo and Group outputs.
- With high-quality digital effects built in, the PMX1609 can deliver a wide range of sound variations even when used on its own. It also includes an EFFECT SEND jack that can be used to connect an external effector.
- The monitor includes a convenient C-R OUT jack. This jack can be used to monitor the main Stereo output, the PFL signal, or the Group 1-2 signals.
- The mixer includes dual AUX SEND jacks and a single RETURN jack. The two independent AUX buses may be used to sends to external effectors and monitor systems.
- Phantom power supply enables easy connection to condenser microphones that run on external power.
- The mixer provides channel-specific INSERT I/O jacks for input channels 1 to 8. These jacks make it possible to insert different effectors into different channels.
- Input channels 1 to 8, 9/10, and 11/12 are each equipped with both an XLR mic input jack and a TRS phone-type line jack. Input channels 13/14 and 15/16 are each equipped with both a TRS line input jack and an RCA line input jack. This wide assortment of connectors enables connection to many different devices, from microphones to line-level devices to stereo-output synthesizers.
- Rack mount brackets included for ultimate flexibility.

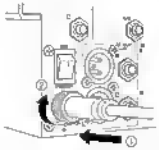
Contents

Before Turning on the Mixer

(1) Be sure that the mixer's power switch is in the STANDBY position.

> Use only the PA-30 adaptor included with this mixer. Use of a different adaptor may result in equipment damage, overheating, or fire.

(2) Connect the power adaptor to the AC ADAPTOR IN connector (①) on the rear of the mixer, and then turn the fastening ring (lock wise (②)) to secure the connection.



(3) Plug the power adaptor into a standard household power outlet.

> - Be sure to unplug the adaptor from the outlet when not using the mixer, or when there are lightning storms in the area.
> - To avoid generating unwanted noise, make sure there is adequate distance between the power adaptor and the mixer.

Turning the Power On

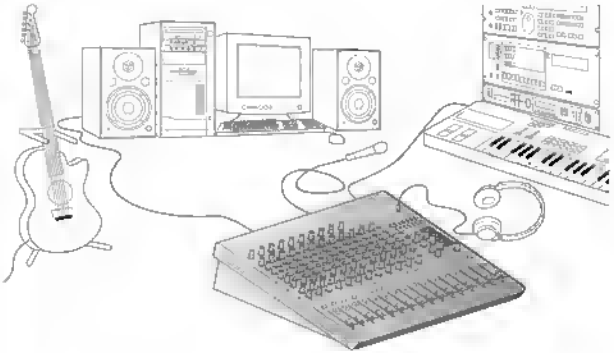
Press the mixer's power switch to the ON position. When you are ready to turn the power off, press the power switch to the STANDBY position.

> Note that trace current continues to flow while the switch is in the STANDBY position. If you do not plan to use the mixer again for a long while, please be sure to unplug the adaptor from the wall outlet.

Making the Most Of Your Mixer

An Introduction

You've got yourself a mixer and now you're ready to use it. Just plug everything in, twiddle the controls, and away you go ... right? Well, if you've done this before you won't have any problems, but if this is the first time you've ever used a mixer you might want to read through this little tutorial and pick up a few basics that will help you get better performance and make better mixes.



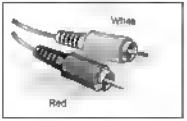
1 A Place For Everything and Everything In Its Place

1-1. A Plethora Of Connectors—What Goes Where?

Questions you're likely to encounter when setting up a system for the first time might include "Why all these different types of connectors on the back of my mixer!" and "What's the difference!".

Let's start by taking a look at the most common connectors types.

The Venerable RCA Pin Jack



This is the "consumer connector," and the one that has been most commonly used on home audio gear for many years. Also known as "phono" jacks (short for "phonogram"), but the term isn't used much these days—besides, it's too easily confusable with "phone" jacks, below. RCA pin jacks are always unbalanced, and generally carry a line-level signal at –10 dB, nominal. You're most likely to use this type of connector when connecting a CD player or other home audio type source to your mixer, or when connecting the output of your mixer to a cassette recorder or similar gear.

The Versatile Phone Jack

The name "phone jack" arose simply because this configuration was first used in telephone switchboards. Phone jacks can be tricky because you can't always tell what type of signal they're designed to handle just by looking at them. It could be unbalanced mono, unbalanced stereo, balanced mono, or an insert patch point. The connector's label will usually tell you what type of signal it handles, as will the owner's manual (you do keep your manuals in a safe place, don't you?). A phone jack that is set up to handle balanced signals is also often referred to as a "TRS" phone jack. "TRS" stands for Tip-Ring-Sleeve, which describes the configuration of the phone plug used.

The Sturdy XLR

This type of connector is generally referred to as "XLR type," and almost always passes a balanced signal. If the corresponding circuitry is designed properly, however, XLR-type connectors will also handle unbalanced signals with no problem. Microphone cables usually have this type of connector, as do the inputs and outputs of most professional audio gear.

1-2. Balanced, Unbalanced—What's the Difference?

In a word: "noise." The whole point of balanced lines is noise rejection, and it's something they're very good at. Any length of wire will act as an antenna to pick up the random electromagnetic radiation we're constantly surrounded by: radio and TV signals as well as spurious electromagnetic noise generated by power lines, motors, electric appliances, computer monitors, and a variety of other sources. The longer the wire, the more noise it is likely to pick up. That's why balanced lines are the best choice for long cable runs. If your "studio" is basically confined to your desktop and all connections are no more than a meter or two in length, then unbalanced lines are fine—unless you're surrounded by extremely high levels of electromagnetic noise. Another place balanced lines are almost always used is in microphone cables. The reason for that is that the output signal from most microphones is very small, so even a tiny amount of noise will be relatively large, and will be amplified to an alarming degree in the mixer's high-gain head amplifier.

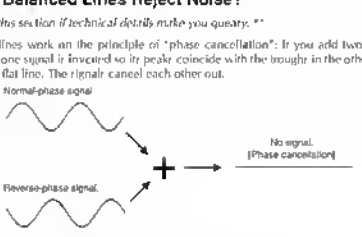
To summarize:

Microphones:	Use balanced lines.
Short line-level runs:	Unbalanced lines are fine if you're in a relatively noise-free environment.
Long line-level runs:	The ambient electromagnetic noise level will be the ultimate deciding factor, but balanced is best.

■ How Do Balanced Lines Reject Noise?

"Skip this section if technical details make you queasy."

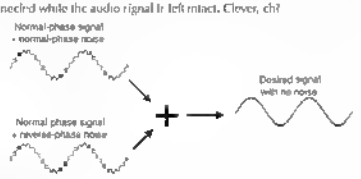
Balanced lines work on the principle of "phase cancellation": If you add two identical signals out of phase (i.e., one signal is inverted so its peaks coincide with the troughs in the other signal), the result is ... nothing. A flat line. The signals cancel each other out.



A balanced cable has three conductors:

1) A ground conductor which carries no signal, just the "ground" or "0" reference against which the signal in the other conductors fluctuates.
2) A "hot" or "+" conductor which carries the normal-phase audio signal.
3) A "cold" or "–" conductor which carries the reverse-phase audio signal.

While the desired audio signals in the hot and cold conductors are out of phase, any noise induced in the line will be exactly the same in both conductors, and thus in phase. The trick is that the phase of one signal is reversed at the receiving end of the line so that the desired audio signals become in-phase, and the induced noise suddenly finds itself out of phase. The out-of-phase noise signal is effectively canceled while the audio signal is left intact. Clever, eh?



1-3. Signal Levels—Decibel Do's and Don'ts

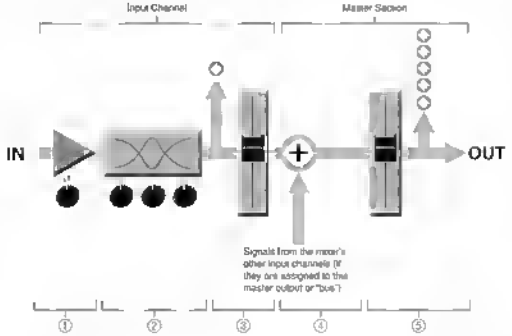
From the moment you start dealing with things audio, you'll have to deal with the term "decibel" and its abbreviation, "dB". Things can get confusing because decibels are a very versatile unit of measure used to describe acoustic sound pressure levels as well as electronic signal levels. To make matters worse there are a number of variations, dBu, dBV, dBm. Fortunately, you don't need to be an expert to make things work. Here are a few basics you should keep in mind.

- "Consumer" gear (such as home audio equipment) usually has line inputs and outputs with a nominal (average) level of –10 dB.
- Professional audio gear usually has line inputs and outputs with a nominal level of +4 dB.
- You should always feed –10 dB inputs with a –10 dB signal. If you feed a +4 dB signal into a –10 dB input you are likely to overload the input.
- You should always feed +4 dB inputs with a +4 dB signal. A –10 dB signal is too small for a +4 dB input, and will result in less-than-optimum performance.
- Many professional and semi-professional devices have level switches on the inputs and/or outputs that let you select –10 or +4 dB. Be sure to set those switches to match the level of the connected equipment.
- Inputs that feature a "Gain" control—such as the mono-channel inputs on your PYLE PRO mixer—will accept a very wide range of input levels because the control can be used to match the input's sensitivity to the signal. More on this later.

2 Where Your Signal Goes Once It's Inside the Box

At first glance the block diagram of even a modest mixer can look like a space-station schematic. In reality, block diagrams are a great aid in understanding how the signal flows in any mixer. Here's a greatly simplified block diagram of a generic mixer to help you become familiar with the way these things work.

2-1. Greatly Simplified Mixer Block Diagram



■ Input Channel

① Head Amp

The very first stage in any mixer, and usually the only stage with significant "gain" or "amplification." The head amp has a "gain" control that adjusts the mixer's input sensitivity to match the level of the source. Small signals (e.g. mics) are amplified, and large signals are attenuated.

② Equalizer

Could be simple bass and treble controls or a full-blown 4-band parametric EQ. When boost is applied the EQ stage also has gain. You can actually overload the input channel by applying too much EQ boost. It's usually better to cut than boost.

③ Channel Peak LED & Fader

The channel peak LED is your most valuable tool for setting the input "gain" control for optimum performance. Note that it is located after the head amp and EQ stage.

■ Master Section

④ Summing Amplifier

This is where the actual "mixing" takes place. Signals from all of the mixer's input channels are "summed" (mixed) together here.

⑤ Master Fader & Level Meter

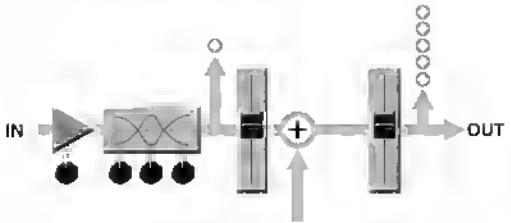
A stereo, mono, or bus master fader and the mixer's main output level meter. There could be several master faders depending on the design of the mixer—i.e. the number of buses or outputs it provides.

3 The First Steps in Achieving Great Sound

Before you even run/ike EQ and effects, or even the overall mix, it is important to make sure that levels are properly set for each individual source. This can't be stressed enough—initial level setup is vitally important for achieving optimum performance from your mixer! Here's why ... and how.

3-1. The Head Amplifier "Gain" Control Is the Key!

Let's review our simplified mixer block diagram:



Each and every "stage" in the mixer's signal path will add a certain amount of noise to the signal: the head amp, the EQ stage, the summing amplifier, and the other buffer and gain stages that exist in the actual mixer circuit (this applies to analog mixers in particular). The thing to keep in mind is that the amount of noise added by each stage is usually not dependent to any significant degree on the level of the audio signal passing through the circuit. This means that the bigger the desired signal, the smaller the added noise will be in relation to it. In tech-speak this gives us a better "signal-to-noise ratio"—often abbreviated as "S/N ratio." All of this leads to the following basic rule:

> ***To achieve the best overall system S/N ratio, amplify the input to the desired average level as early as possible in the signal path.***

In our mixer, that means the head amplifier. If you don't get the signal up to the desired level at the head amplifier stage, you will need to apply more gain at later stages, which will only amplify the noise contributed by the preceding stages. Just remember that too much initial gain is bad too, because it will overload our channel circuitry and cause clipping.

3-2. Level Setup Procedure For Optimum Performance

Now that we know what we have to do, how do we do it? If you like another quick look at the mixer block diagram you'll notice that there's a peak indicator located right after the head amplifier and EQ stages, and therein lays our answer! Although the exact procedure you use will depend on the type of mixer you use and the application, as well as your personal preferences, here's a general outline:

1 Start by setting all level controls to their minimum: master faders, group faders (if provided), channel faders, and input gain controls. Also make sure that no EQ is applied (no boost or cut), and that all effects and dynamic processors included in the system are defeated or bypassed.

2 Apply the source signal to each channel one at a time: have singers sing, players play, and playback devices play back at the loudest expected level. Gradually turn up the input gain control while the signal is being applied to the corresponding channel until the peak indicator begins to flash, then back off a little so that the peak indicator flashes only occasionally. Repeat for each active channel.

3 Raise your master fader(s)—and group faders if available—to their nominal levels (this will be the "0" markings on the fader scale).

4 Now, with all sources playing, you can raise the channel faders and set up an initial rough mix.

That's basically all there is to it. But do keep your eyes on the main output level meters while setting up the mix to be sure you don't stay in the "peak zone" all the time. If the output level meters are peaking constantly you will need to lower the channel faders until the overall program falls within a good range—and this will depend on the "dynamic range" of your program material.

4 External Effects, Monitor Mixes, and Gronps

4-1. AUX Buses For Monitor Sends and Overall Effects

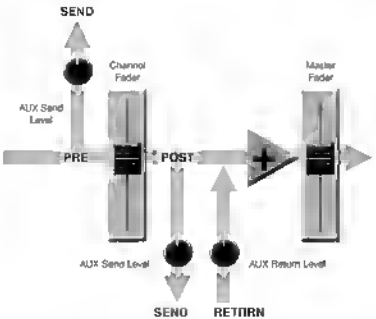
There are a number of reasons why you might want to "tap" the signal flowing through your mixer at some point before the main outputs, the two most common being 1) to create a monitor mix that is separate from the main mix, and 2) to process the signal via an external effect unit and then bring it back into the mix. Both of these functions, and more, can be handled by the mixer's AUX (Auxiliary) buses and level controls. If the mixer has two AUX buses, then it can handle both functions at the same time. Larger mixing consoles can have 6, 8, or even more auxiliary buses to handle a variety of monitoring and processing needs.

Using the AUX buses and level controls is pretty straightforward. The only thing you need to consider is whether you need a "pre-fader" or "post-fader" send. AUX sends often feature a switch that allows you to configure them for pre- or post-fader operation.

Pre/Post—What's the difference?

pre	post
A "pre-fader" signal is taken from a point before the channel fader, so the send level is affected only by the AUX send level control and not by the channel fader Pre fader sends are most commonly used to provide monitor mixes	A "post-fader" signal is taken from a point after the channel fader, so its level will be affected by both the AUX send level control and the channel fader. Post fader sends are most commonly used in conjunction with the mixer's AUX or effect returns for external effect processing

Pre-fader send for a monitor mix. The send signal is fed to the monitor power amplifier and speaker system. The channel fader does not affect the send level so the monitor mix remains independent of the main mix. No return signal is used in this case

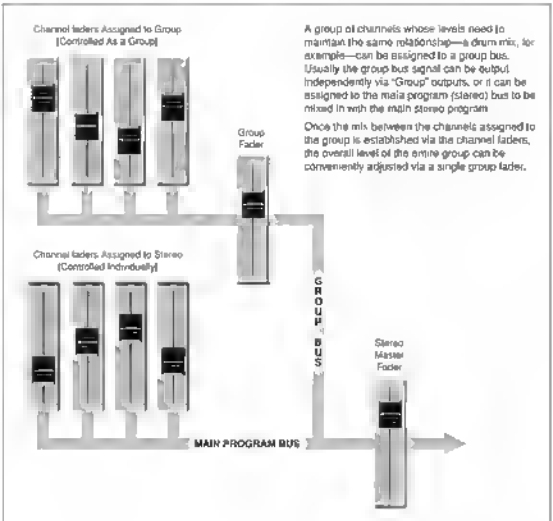


Post-fader send for external effects processing. The send signal is fed to the external effect unit—a reverb unit, for example—and the output from the effect unit is returned to the AUX Return jack and mixed back into the main program. The send level is affected by the channel fader so the effect level always remains in proportion to the channel signal.

4-2. Using Groups

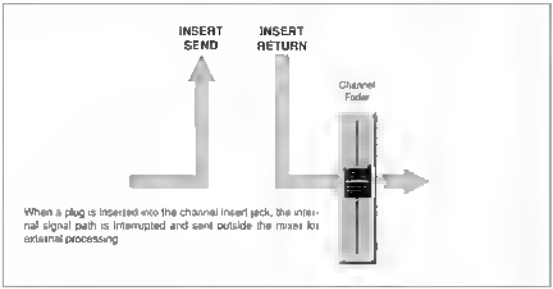
Group buses and faders can greatly simplify the mixing process—particularly in live situations in which changes have to be made as quickly as possible. If you have a group of channels that need to be adjusted all together while maintaining their relative levels, grouping is the way to go. Simply assign the group to a group bus, and make sure that group is also assigned to the main program bus. Then you can adjust the overall level of the group using a single group fader, rather than having to attempt to control multiple channels faders simultaneously.

Group buses usually also have their own outputs, so you can send the group signal to a different external destination from the main mix.

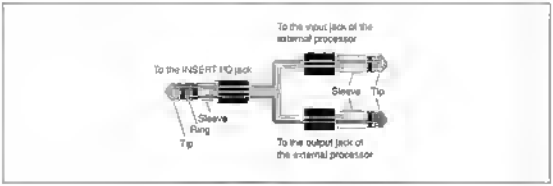


4-3. Channel Inserts for Channel-specific Processing

Another way to get the mixer's signal outside the box is to use the channel inserts. The channel inserts are almost always located before the channel fader and, when used, actually "break" the mixer's internal signal path. Unlike the AUX sends and returns, the channel insert only applies to the corresponding channel. Channel inserts are most commonly used for applying a dynamics processor such as a compressor or limiter to a specific channel—although they can be used with just about any type of in/out processor.



Channel insert jacks must be used with a special insert cable that has a TRS phone jack on one end and mono phone jacks on the split "Y" end. One of the mono phone jacks carries the "send" signal to be fed to the input of the external processor, and the other carries the "return" signal from the output of the processor.



5 Making Better Mixes

5-1. Approaching the Mix—Where Do You Start?

Mixing is easy, right? Just move the faders around until it sounds right? Well, you can do it that way, but a more systematic approach that is suited to the material you're mixing will produce much better results, and faster. There are no rules, and you'll probably end up developing a system that works best for you. But the key is to *develop a system* rather than working haphazardly. Here are a few ideas to get you started:

Faders Down

It might sound overly simple, but it is usually a good idea to start with all channel faders off—all the way down. It's also possible to start with all faders at their nominal settings, but it's too easy to lose perspective with this approach. Start with all faders down, then bring them up one by one to fill out the mix. But which channel should you start with?

Example1:
Vocal Ballad Backed by Piano Trio

What are you mixing? Is it a song in which the vocals are the most important element? If so you might want to build the mix around the vocals. This means bringing the vocal channel up to nominal first in your level setup procedure has been done properly this will be a good starting point), and then adding the other instruments. What you add next will depend on the type of material you are working with and your approach to it. If the vocals are backed by a piano trio and the song is a ballad, for example, you might want to bring in the piano next (and get the vocal/piano relationship just right), then bring in the bass and drums to support the overall sound.

Example2:
Funky R&B Groove

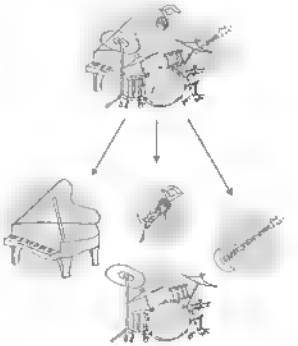
The approach will be totally different if you're mixing a funky R&B number that centers on the groove. In this case most engineers will start with the drums, and then add the bass. The relationship between the drums and bass is extremely important to achieve the "drive" or groove the music rides on. Pay particular attention to how the bass works with the kick (bass drum). They should almost sound like a single instrument—with the kick supplying the punch and the bass supplying the pitch. Once again, there are no rules, but these are concepts that have been proven to work well.

Music First—Then Mix

In any case, the music comes first. Think about the music and let it guide the mix, rather than trying to do things the other way around. What is the music saying and what instrument or technique is being used to drive the message? That's where the focus of your mix should be. You're using a high-tech tool to do the mixing, but the mix itself is as much art as the music. Approach it that way and your mixes will become a vital part of the music.

5-2. Panning For Cleaner Mixes

Not only does the way you pan your individual channels determine where the instruments appear in the stereo sound field, but it is also vital to give each instrument it's own "space" so that it doesn't conflict with other instruments. Unlike live sound in a real acoustic space, recorded stereo sound is basically 2-dimensional (although some types of surround sound are actually very 3-dimensional), and instruments positioned right on top of each other will often get in each other's way—particularly if they are in the same frequency range or have a similar sound.



Spread them Out!

Position your instruments so they have room to "breathe," and connect in the most musical way with other instruments. Sometimes, however, you'll want to deliberately pan sounds close together, or even right on top of one another, to emphasize their relationship. There are no hard-and-fast rules. Normally (but this is not a rule), bass and lead vocals will be panned to center, as will the kick drum if the drums are in stereo.

5-3. To EQ Or Not To EQ

In general, less is better. There are many situations in which you'll need to cut certain frequency ranges, but use boost sparingly, and with caution. Proper use of EQ can eliminate interference between instruments in a mix and give the overall sound better definition. Bad EQ—and most commonly bad boost—just sounds terrible.

Cut For a Cleaner Mix

For example: cymbals have a lot of energy in the mid and low frequency ranges that you don't really perceive as musical sound, but which can interfere with the clarity of other instruments in these ranges. You can basically turn the low EQ on cymbal channels all the way down without changing the way they sound in the mix. You'll hear the difference, however, in the way the mix sounds: more "spacious," and instruments in the lower ranges will have better definition. Surprisingly enough, piano also has an incredibly powerful low end that can benefit from a bit of low-frequency roll-off to let other instruments—notably drums and bass—do their jobs more effectively. Naturally you won't want to do this if the piano is playing solo.

The reverse applies to kick drums and bass guitars: you can often roll off the high end to create more space in the mix without compromising the character of the instruments. You'll have to use your ears, though, because each instrument is different and sometimes you'll want the "snap" of a bass guitar, for example, to come through.

Boost With Caution

If you're trying to create special or unusual effects, go ahead and boost away as much as you like. But if you're just trying to achieve a good-sounding mix, boost only in very small increments. A tiny boost in the midrange can give vocals more presence, or a touch of high boost can give certain instruments more "air." Electric and strings don't sound clear and clean by trying to cut to remove frequencies that are cluttering up the mix rather than trying to boost the mix into clarity.

One of the biggest problems with too much boost is that it adds gain to the signal, increasing noise and potentially overloading the subsequent circuitry.

5-4. Ambience

Judicious application of reverb and/or delay via the mixer's AUX busses can really polish a mix, but too much can "wash out" the mix and reduce overall clarity. The way you set up your reverb sound can make a huge difference in the way it meshes with the mix.

Reverb/Delay Time

Different reverb/delay units offer different capabilities, but most offer some means of adjusting the reverb time. A little extra time spent matching the reverb time to the music being mixed can mean the difference between great and merely average sound. The reverb time you choose will depend to a great degree on the tempo and "density" of the mix at hand. Slower tempos and lower densities (i.e. sparser mixes with less sonic activity) can sound good with relatively long reverb times. But long reverb times can completely wash out a busier more active piece of music. Similar principles applies to delay.

Reverb Tone

How "bright" or "hazy" a reverb sound is also has a large impact on the sound of your mix. Different reverb units offer different means of controlling this—balance between the high- and low-frequency reverb times, simple EQ, and others. A reverb that is too bright will not only sound unnatural, but it will probably get in the way of the other highs you want to come through in your mix. If you find yourself hearing more high-end reverb than mix detail, try reducing the brightness of the reverb sound. This will allow you to get full-bodied ambience without compromising clarity.

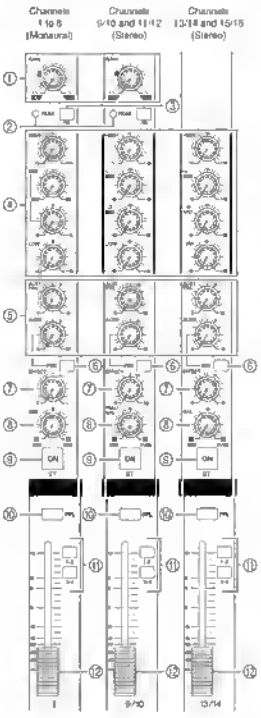
Reverb Level

It's amazing how quickly your ears can lose perspective and fool you into believing that a really washed-out mix sounds perfectly fine. To avoid falling into this trap start with reverb level all the way down, then gradually bring the reverb into the mix until you can just hear the difference. Any more than this normally becomes a "special effect." You don't want reverb to dominate the mix unless you are trying to mimic the effect of a band in a cave—which is a perfectly legitimate creative goal if that's the sort of thing you're aiming for.

5-5. Built-in Effects & EQ

Your PMX mixer features a high-performance internal effect system and graphic equalizer that offers extraordinary sound-processing power and versatility without the need for external equipment. The internal DSP (Digital Signal Processor) lets you individually add reverb and delay to each channel in the same way that you can with an external effect unit – but you don't need to wire up any extra gear, and won't suffer the signal quality loss that external connections sometimes entail. The graphic equalizer is ideal for shaping the response of the overall mix, and for minimizing feedback in live situations. For details see page 22.

Channel Control Section



① **GAIN Control**

Adjusts the input signal level. To get the best balance between the S/N ratio and the dynamic range, adjust the level so that the peak indicator (②) comes on only at about maximum input level.

The –60 to –10 scale indicates the MIC-input adjustment level. The –34 to +10 scale indicates the LINE-input adjustment level.

② **PEAK Indicator**

Detects the peak level of the post-equalizer signal, and lights up red when the level reaches 3 dB below the clipping level. On stereo input channels equipped with XLR jacks (CHs 9/10 and 11/12), detects both the post-equalizer and post-mic-amp peak levels, and lights up red if either of these levels reaches 3 dB below the clipping level.

③ **/80 Switch (High Pass Filter)**

This switch toggles the HPF on or off. To turn the HPF on, press the switch in (▬). The HPF cuts frequencies below 80 Hz. (But note that regardless of the switch setting, the mic or line inputs are always fed through the HPF.)

④ **Equalizer**

- Monaural (CHs 1 to 8)

This three-band equalizer adjusts the channel's high, mid, and low frequency bands. Setting the knob to the ▼ position produces a flat frequency response. Turning the knob to the right boosts the corresponding frequency band, while turning to the left cuts the band. The following table shows the EQ type, base frequency, and maximum cut/boost for each of the three bands.

Band	Type	Base Frequency	Maximum Cut/Boost
HIGH	Shelving	10 kHz	±15 dB
MID	Peaking	250 Hz – 5 kHz (variable)	
LOW	Shelving	100 Hz	

- Stereo channels (CHs 9/10, 11/12, 13/14, 15/16)

This four-band equalizer adjusts the channel's high, hi-mid, lo-mid, and low frequency bands. Setting the knob to the ▼ position produces a flat frequency response. Turning the knob to the right boosts the corresponding frequency band, while turning to the left cuts the band. The following table shows the EQ type, base frequency, and maximum cut/boost for each of the four bands.

Band	Type	Base Frequency	Maximum Cut/Boost
HIGH	Shelving	10 kHz	±15 dB
HI-MID	Peaking	3 kHz	
LO MID	Peaking	800 Hz	
LOW	Shelving	100 Hz	

⑮ AUX1 and AUX2 Controls

The AUX1 knob controls the signal level that the channel sends to the AUX1 bus; the AUX1 knob controls the signal level to the AUX2 bus. These knobs should generally be set close to the ▼ position.

If you are using stereo channels, the signals from the L (odd) and R (even) channels are mixed and sent to the AUX1 and AUX2 buses.

NOTE

These controls allow you to output the signal to the AUX buses regardless of the setting of the ST switch ⑭.

⑯ PRE Switch

Selects whether the pre-fader or the post-fader signal is fed to the AUX2 bus. If you set the switch on (▬), the mixer sends the pre-fader signal—the signal prior to passage through channel fader ⑫—to the AUX2 bus, so that AUX2 output is not affected by the fader. If you set the switch off (▬), the mixer sends the post-fader signal to the AUX2 bus.

Note that this switch applies to AUX2 only. The signal to the AUX1 bus always passes through the channel fader first.

⑰ EFFECT Control

Adjusts the level of the signal sent from the channel to the EFFECT bus. Note that the signal level will also vary according to the setting of the channel fader.

If you are using stereo channels (CH1 & 2/3, 4/5, 6/7, 8/9, 10/11, 12/13, 14/15, 16/17), the signals from the L (odd) and R (even) channels are mixed and then sent to the EFFECT bus.

⑱ PAN Control (CHs 1 to 8)
PAN/BAL Control (9/10 and 11/12)
BAL Control (13/14 and 15/16)

The PAN control determines the positioning of the channel's signal on the Group 1-2/3-4 buses or on the Stereo L and R buses.

The BAL control knob sets the balance between left and right channels. Signals sent to the L input (odd channel) feed to the Group 1/3 bus or to the Stereo L bus; signals sent to the R input (even channel) feed to the Group 2/4 bus or the Stereo R bus.

NOTE

On channels where the knob provides both PAN and BAL controls (9/10 and 11/12), the knob operates as a PAN control if you are inputting through the MIC jack or into the L (MONO) input only, and operates as a BAL control if you are inputting into both L and R inputs.

⑲ ST Switch

This switch assigns the channel's signal to the Stereo L and R buses. To send the signal to the Stereo bus, set the switch on by pressing it in (▬). The switch lights up orange to indicate that it is on.

⑳ PEL (Pre-Fader Listen) Switch

This switch lets you monitor the channel's pre-fader signal. To set the switch on, press it in (▬) so that it lights up. When the switch is on, the mixer outputs the channel's pre-fader signal to the PHONES and CR OUT jacks for monitoring.

㉑ GROUP Switches

Use these switches to send the channel's signal to the Group 1-2 and/or Group 3-4 buses. Setting the switch on (▬) causes the signal to be sent to the corresponding group buses.

NOTE

These switches allow you to assign the signal to either or both groups regardless of the setting of the ST switch ⑲.

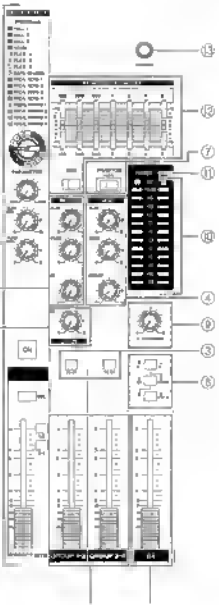
㉒ Channel Fader

Adjusts the output level of the signal being input to the channel. Use these faders to adjust the volume balance among the various channels.

NOTE

To reduce noise, set the fader sliders for unused channels all the way down.

Master Control Section



③ TO ST Switch

If this switch is on (■), the mixer sends the signals processed by the GROUP faders (1-2, 3-4) onto the Stereo bus. The Group 1/3 signal go to Stereo L, and the Group 2/4 signal go to Stereo R.

④ Master SEND

- **AUX1 and AUX2 Controls**
  Adjust the level of the signal output to the AUX1 SEND and AUX2 SEND jacks.
- **Master EFFECT Control**
  Adjusts the level of the signal on the EFFECT bus. This is the signal that is output through the EFFECT jack.

NOTE

These Master SEND controls do not affect the level of the signal sent from the internal digital effector onto the Master EFFECT bus.

⑤ RETURN (AUX1, AUX2, and ST Controls)

- **AUX1 and AUX2 Controls**
  Adjust the level of the mixed L/R signal sent from the RETURN jacks (L (MONO) and R) to the AUX1 and AUX2 buses.
- **ST Control**
  Adjusts the level of the signal sent from the RETURN jacks (L (MONO) and R) to the Stereo bus.

NOTE

If you supply a signal to the RETURN L (MONO) jack only, the mixer outputs the identical signal to both the L and R Stereo buses.

⑥ 2TR IN Control

Adjusts the level of the signal sent from the 2TR IN jack to the Stereo bus.

⑦ PHANTOM +48 V Switch

This switch toggles phantom power on and off. If you set the switch on, the mixer supplies power to all channels that provide MIC input jacks (CH= 1-8, 9/10, 11/12). Set this switch on (■) when using one or more condenser microphones.

NOTE

When this switch is on, the mixer supplies DC +48 V power to pins 2 and 3 of all XLR-type jacks.

- Be sure to leave this switch off (□) when you are not using phantom power. Humming or damage may result if you connect to an unbalanced device or to an ungrounded customer while this switch is on (■). But note that the switch may be left on (■) without problem when connecting to balanced electronic microphones.
- To avoid damage to speakers, be sure to turn all amplifiers for powered speakers) before turning this switch on or off.

① ST Master Fader

Adjusts the signal level to the ST OUT jacks.

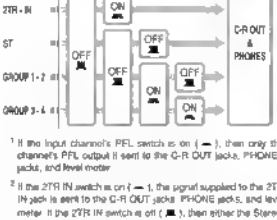
② GROUP Faders (1-2, 3-4)

Adjust the signal level to the GROUP OUT 1 to 4 jacks.

⑨ Level-Meter Signal Switches

These level-meter switches, together with the channel PFL switches, select the signal that is sent through the C-R/PHONES control to the C-R OUT jacks, the PHONES jack, and the level meter.

The following illustration shows how the switch settings correspond to the signal selection.



[1] If the input channel's PFL switch is on (▬), then only the channel's PFL output is sent to the C-R OUT jacks, PHONES jacks, and level meter.

[2] If the 2TR IN switch is on (▬), the signal supplied to the 2TR IN jack is sent to the C-R OUT jacks, PHONE jacks, and level meter. If the 2TR IN switch is off (▄), then either the Stereo, Group 1-2, or Group 3-4 signal is sent to the C-R OUT jacks (as determined by the ST-GROUP and GROUP toggle switches).

⑩ C-R/PHONES Control

Controls the level of the signal output to the PHONES jack and the C-R L and R jacks.

⑪ Level Meter

This LED display shows the level of the signal selected by the selection switches described in ⑨, above (the level in the C-R OUT and PHONES jacks). The "0" point corresponds to the standard output level. The indicator lights up red when the output has the clipping level.

⑫ POWER Indicator

This indicator lights up when the mixer's power is ON.

⑬ ST GRAPHIC EQUALIZER

This 7-band equalizer adjusts the sound of the signal sent to the ST OUT jacks. The equalizer is effective only if the GEQ switch is set on (▬). The equalizer cuts or boosts each band (125, 250, 500, 1k, 2k, 4k, and 8k Hz) over a range of ±12 dB.

⑭ PHONES Jack

Connector for headphones. This is a balanced stereo phone-type output jack.

NOTE

The signal monitored by these jacks is selected by the Level Meter Signal switches and the channel PFL switches.

⑮ DIGITAL EFFECT

- PROGRAM Dial

Selects the internal digital effect to be applied. You can select from 16 effects, as shown in the table.

1	HALL 1	9	VOCAL ECHO 1
2	HALL 2	10	VOCAL ECHO 2
3	HALL 3	11	VOCAL ECHO 3
4	ROOM	12	VOCAL ECHO 4
5	PLATE 1	13	VOCAL REVERB 1
6	PLATE 2	14	VOCAL REVERB 2
7	PLATE 3	15	VOCAL REVERB 3
8	GATE REVERB	16	VOCAL REVERB 4

- PARAMETER Control

Adjusts the parameter (depth, speed, etc.) for the selected effect.

- AUX1 and AUX2 Controls

Adjust the level of the signal sent from the internal digital effector to the AUX1 and AUX2 buses.

- ON Switch

Switches use of the internal effect on or off. The internal effect is applied only if this switch is turned on (▬).

- PFL Switch

Set this switch on (▬) if you wish to output the effect signal to the PFL bus.

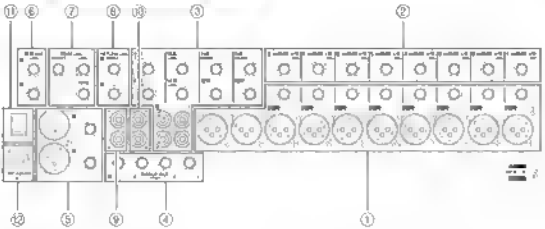
- GROUP Switches (1-2, 3-4)

Set one or both switches on (▬) to output the internal effect signal to the GROUP 1-2 and/or GROUP 3-4 buses.

- EFFECT RTN Fader

Adjusts the signal level from the internal digital effector to the ST/STEREO bus.

Rear Input/Output Section



① Channel Input Jacks

- MIC jacks (CHs 1 to 8, 9/10, 11/12)
  These are balanced XLR-type input jacks.
- LINE jacks (CHs 1 to 8)
  These are balanced phone-type input jacks. You can connect either balanced or unbalanced phone plugs to these jacks.

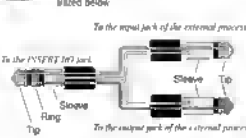
NOTE

Where an input channel provides both a MIC jack and a LINE jack, you may use either one of these jacks but you may not use both at the same time. Please connect to only one of these jacks on each channel.

⑦ INSERT I/O Jacks

These are unbalanced phone-type input/output jacks. Each of these jacks is positioned between the equalizer and fader of the corresponding input channel (CHs 1 to 8). These jacks can be used to independently connect these channels to devices such as graphic equalizers, compressors, and noise filters. These are TRS (tip, ring, sleeve) phone jacks that support balanced operation.

Connection to an INSERT I/O jack requires a special separately sold insertion cable such as illustrated below.



The signal output from the INSERT I/O jacks is reverse-phased. This will not be a problem in connecting the jack to an effector. If using the jack to output to an external device, however, please be aware of possible phase conflicts with other signals.

③ Channel Input Jacks

These are unbalanced input jacks. Two jack types are provided: phone type (CHs 9/10 to 15/16) and RCA pin type (CHs 13/14, 15/16). Use these jacks to input stereo signals, inputting the L signal(s) to the odd-numbered channel(s) and the R signal(s) to the even-numbered channel(s).

NOTE

- Where a channel provides both a phone jack and an RCA pin jack, you may use either one of these jacks but you may not use both at the same time. Please connect to only one of these jacks on each channel.
- The phone-type jacks for CH9/10 and 11/12 also support monaural input. Specifically, if you input only into the L(MONO) jack of either pair (while leaving the R jack empty), the mixer will propagate the same signal through both the L(MONO) and R inputs.

④ GROUP OUT (1 to 4) Jacks

These are impedance-balanced phone-type output jacks that output the Group 1-2/3-4 signals. Use these jacks to connect to the input jacks of an MTR, external mixer, or other such device.

⑤ ST OUT (L, R) Jacks

These jacks output the mixed signal whose level is adjusted by the ST fader in the Master Control section. Output is in stereo (L and R). You use these jacks, for example, to connect to the power amplifier driving your main speakers.

- XLR jacks
  XLR-type balanced output jacks
- Phone jacks
  TRS phone-type balanced output jacks

⑥ C-R OUT Jacks

These are impedance-balanced phone-type output jacks. These jacks output the mixed signal whose level is adjusted by the C-R/PHONES control. Output is in stereo (L and R). These jacks are typically used to connect to a monitor system.

NOTE

The signal monitored by these jacks is selected by the Level Meter Signal switches and the channel PFL switches.

⑦ SEND Jacks

- AUX1, AUX2

These are impedance-balanced phone-type output jacks. These jacks output the signal from AUX1 and AUX2, respectively. You use these jacks, for example, to connect to an effector or to a cue box or other such monitoring system.

- EFFECT

This is an impedance-balanced phone-type output jack that outputs the signal from the EFFECT bus. You use this jack, for example, to connect to an external effector.

⑧ RETURN L (MONO), R Jacks

These are unbalanced phone-type input jacks. The signal received by these jacks is sent to the Stereo bus and the AUX1 and AUX2 buses. These jacks are typically used to receive a return signal from an external effector (reverb, delay, etc.).

NOTE

These jacks can also be used as an auxiliary stereo input. If you connect to the L(MONO) jack only, the mixer will recognize the signal as monaural and will propagate the identical signal on both L and R jacks.

⑨ REC OUT (L, R) Jacks

These are unbalanced RCA-pin-type output jacks. By connecting these jacks to an external recorder, you can record the same signal that is being output from the ST OUT jacks.

NOTE

The mixer's ST Master Fader does not operate on the signal output from these jacks. Be sure to make appropriate level adjustments at the external recording device.

⑩ 2TR IN Jacks

These unbalanced RCA-pin-type input jacks are used to input a stereo sound source. Use these jacks when you want to connect a stereo sound source (CD or DAT) directly to the mixer for monitoring.

NOTE

You can adjust the signal level using the 2TR IN control in the Master Control section.

⑪ POWER Switch

Use this switch to set the mixer power to ON or STANDBY.

⚠

Note that some current continues to flow while the switch is in the STANDBY position. If you do not plan to use the mixer again for a long while, be sure to unplug the adapter from the wall outlet.

⑫ AC ADAPTOR IN Connector

Connects to the included PA-30 power adaptor (see page 5).

⚠

Use only the PA-30 adaptor included with this unit. Use of a different adaptor may result in fire or electric shock.

Connector Polarities

MIC INPUT, ST OUT	Pin 1: Ground Pin 2: Hot (+) Pin 3: Cold (–)	INPUT OUTPUT 
LINE INPUT (monaural channels), GROUP OUT, ST OUT, C-R OUT, AUX1, AUX2, EFFECT*	Tip: Hot (+) Ring: Cold (–) Sleeve: Ground	
INSERT I/O	Tip: Output Ring: Input Sleeve: Ground	
PHONES	Tip: L Ring: R Sleeve: Ground	
RETURN, LINE INPUT (stereo channels)	Tip: Hot Sleeve: Ground	

* These jacks will also accept connection to monaural phone plugs. If you use monaural plugs, the connection will be unbalanced.

Setup Procedure

(1) Before connecting to microphones and instruments, be sure that all devices are turned off. Also be sure that all of the mixer's channel faders and master control faders are set all the way down.

(2) For each connection, connect one end of the cable to the external microphone or instrument and connect the other end to the appropriate LINE or MIC jack on the mixer.
(LINE jacks: CH-1 to 8, MIC jacks: CH-1 to 8, 9/10, 11/12)

NOTE

Where an input channel provides both a MIC jack and a LINE jack, you may use either one of these jacks but you may not use both at the same time. Please connect to only one of these jacks on each channel.

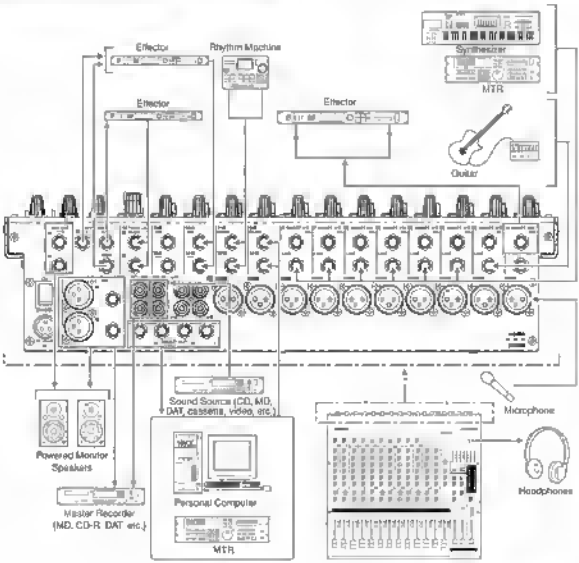
(3) To avoid causing damage to speakers, power up the devices in the following order: Peripheral devices → mixer → power amps (or powered speakers).

NOTE

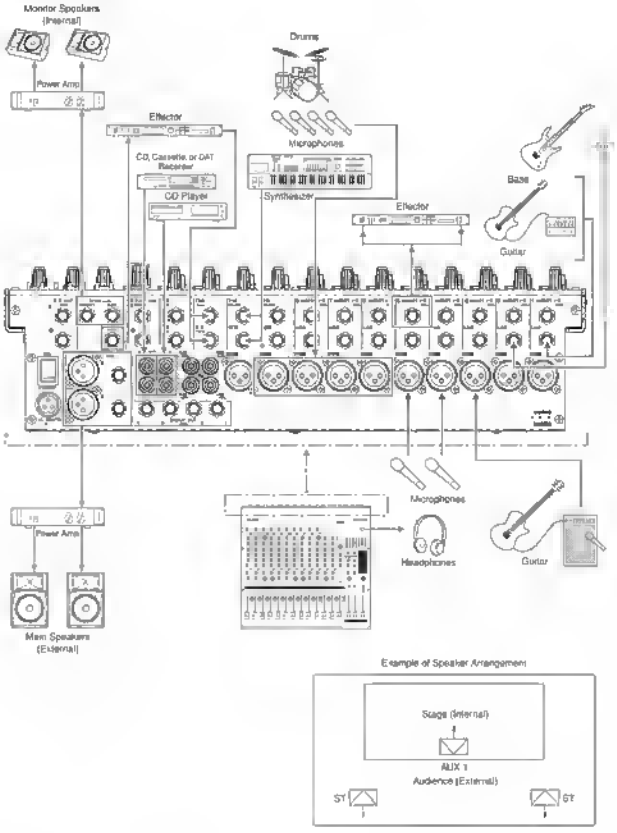
When shutting the system down, turn off the power in the opposite order. Power amps (powered speakers) → mixer → peripheral devices

Setup Examples

■ **Home Recording**



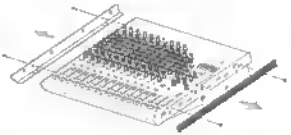
■ **Sound Reinforcement for Live Performance**



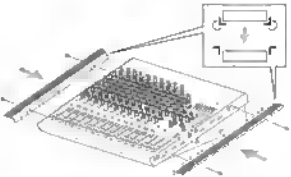
Rack Mounting

Mounting

(1) Two metal rack-mount supports are screwed onto the unit. Use a screwdriver to remove these supports.



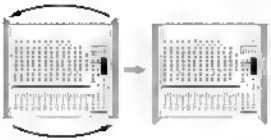
(2) Turn the supports over, and fasten them into place again using the same screws.



(3) Mount the unit into the rack, and fasten it into place.

NOTE

If you wish you may move the left support to the right side and the right support to the left side, as shown in the drawing.



Do not install the mixer near power amps or other heat-generating devices.

Specifications

■ General Specifications

Frequency Characteristics (ST OUT)	20 Hz–20 kHz +1 dB, –3 dB @+4 dBu, 600 Ω (with gain control at maximum level)	
Total Harmonic Distortion (ST OUT)	0.1% (THD+N) @+14 dBu, 20 Hz–20 kHz, 600 Ω (with gain control at maximum level)	
Hum and Noise [1]	–128 dBu	Equivalent input noise (CHs 1 to 8)
	–100 dBu	Residual output noise (ST OUT)
	–88 dBu (92 dB S/N)	ST, GROUP Master fader at nominal level, all Ch assign switches off
	–81 dBu (85 dB S/N)	AUX, EFFECT master control at nominal level, all channel mix controls at minimum level
	–84 dBu (88 dB S/N)	ST, GROUP Master fader and one Ch fader at nominal level. (CHs 1 to 8)
Maximum Voltage Gain [2]	66 dB CH MIC INPUT → CH INSERT OUT 84 dB CH MIC INPUT → GROUP OUT/ST OUT (CH to ST) 94 dB CH MIC INPUT → ST OUT (GROUP to ST) 82.2 dB CH MIC INPUT → REC OUT (CH to ST) 76 dB CH MIC INPUT → AUX SEND (PRE) 96 dB CH MIC INPUT → AUX SEND (POST)/EFFECT SEND 58 dB CH LINE INPUT → GROUP OUT/ST OUT (CH to ST) 84 dB ST CH MIC INPUT → GROUP OUT/ST OUT (CH to ST) 58 dB ST CH LINE INPUT → GROUP OUT/ST OUT (ST to ST) 47 dB ST CH LINE INPUT → AUX SEND (PRE) 57 dB ST CH LINE INPUT → AUX SEND (POST)/EFFECT SEND 34 dB ST CH INPUT → GROUP OUT/ST OUT (ST to ST) 16 dB AUX RETURN → ST OUT 9 dB AUX RETURN → AUX SEND 27.5 dB 2TR INPUT → ST OUT	
Monaural/Stereo Input Gain Control	44 dB variable	
Monaural/Stereo High Pass Filter	80 Hz 12 dB/octave	
Crosstalk (1 kHz)	- 70 dB between input channels –70 dB between input/output channels (CH INPUT)	
Monaural Input Channel Equalization Max. Variation (CHs 1 to 8) [3]	±15 dB HIGH 10 kHz shelving MID 0.25–5 kHz peaking LOW 100 Hz shelving	
Stereo Input Channel Equalization Max. Variation (CHs 9/10 to 15/16) [3]	±15 dB HIGH 10 kHz shelving HI-MID 3 kHz peaking LO-MID 800 Hz peaking LOW 100 Hz shelving	
Graphic Equalizer	7 band (125, 250, 500 1 k, 2 k, 4 k, 8 k) Max. Variation ±12 dB	
Internal Digital Effects	16 programs, parameter control	
Monaural/Stereo Input Peak Indicator	On each channel red indicator lights if post EQ signal (on ST channels, if either post-EQ signal or post-mic-amp signal) comes within 3 dB of the clipping level	
Level Meters	Two 12-point LED meters (Stereo (L, R)) Peak point, red indicator +5, +3, +1, 0 and 6 points, yellow indicators –1, –3, –5, –7, –10, –15, –20: green indicators	
Phantom +48 VDC Power (Balanced Input)	Supplied when Phantom +48 V switch is ON	
Included Accessory	Power adaptor (PA-30)	
Power Supply	AC 110V/220V Adaptor	
Power Consumption	51 W	
Max. Dimensions (W × H × D)	423 × 108 × 410 mm	
Weight	5.5 kg	

Where 0 dBu = 0.775 V and 0 dBV = 1 V

[1] Rs = 150 ohms
Measured with a 2.7 kHz, –6 dB/oct low pass filter (equivalent to 20 kHz, — filter).
(CH MIC INPUT to ST GROUP OUT/AUX EFFECT SEND)

[2] Turning PAN/BAL to left or right.

[3] Shelving turnover/falloff frequency: 3 dB before maximum cut or boost.

■ Input Specifications

Input Connector	Gain	Input Impedance	Appropriate Impedance	Sensitivity*	Rated Level	Max. Before Clipping	Connector Specifications
MIC INPUT (CH# 1 to 8)	-60	3 kΩ	50~600 Ω mic	-60 dBu (0.078 mV)	-60 dBu (0.775 mV)	-40 dBu (7.75 mV)	XLR-3-31 type (balanced)
	+16			-36 dBu (12.3 mV)	+16 dBu (12.3 mV)	+44 dBu (123 V)	
LINE INPUT (CH# 1 to 8)	-34	10 kΩ	600 Ω line	-54 dBu (1.55 mV)	-34 dBu (15.5 mV)	-14 dBu (155 mV)	Phone jack (TRS) (balanced [T: hot, R: cold, S: ground])
	+10			+10 dBu (245 mV)	+10 dBu (2.45 V)	+30 dBu (24.5 V)	
ST CH MIC INPUT (CH9(L)/CH10(R), CH11(L)/CH12(R))	-60	3 kΩ	50~600 Ω mic	-80 dBu (0.078 mV)	-60 dBu (0.775 mV)	-40 dBu (7.75 mV)	XLR-3-31 type (balanced)
	-16			-36 dBu (12.3 mV)	-16 dBu (123 mV)	-10 dBu (245 mV)	
ST CH LINE INPUT (CH9(L)/CH10(R), CH11(L)/CH12(R))	-34	10 kΩ	600 Ω line	-54 dBu (1.55 mV)	-34 dBu (15.5 mV)	-14 dBu (155 mV)	Phone jack (unbalanced)
	+10			+10 dBu (245 mV)	+10 dBu (2.45 V)	+30 dBu (24.5 V)	
ST CH. INPUT (CH13(L)/CH14(R), CH15(L)/CH16(R))		10 kΩ	600 Ω line	-30 dBu (24.5 mV)	-10 dBu (245 mV)	+10 dBu (2.45 V)	Phone jack (unbalanced); RCA pin jack
CH INSERT IN (CH# 1 to 8)		10 kΩ	600 Ω line	-20 dBu (77.5 mV)	0 dBu (0.775 V)	+20 dBu (7.75 V)	Phone jack (TRS) (unbalanced [T: out, R: in, S: ground])
AUX RETURN (L, R)		10 kΩ	600 Ω line	-12 dBu (195 mV)	+4 dBu (1.23 V)	+24 dBu (12.3 V)	Phone jack (TRS) (unbalanced [T: hot, S: ground])
2TR IN (L, R)		10 kΩ	600 Ω line	-20 dBV (50.1 mV)	-10 dBV (316 mV)	+10 dBV (3.16 V)	RCA pin jack

When 0 dBu = 0.775 V and 0 dBV= 1 V

* Input sensitivity: the lowest level that will produce the nominal output level when the unit is set to maximum gain.

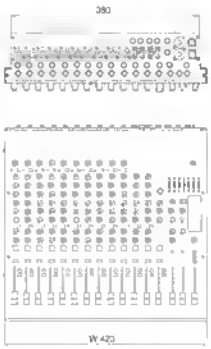
■ Output Specifications

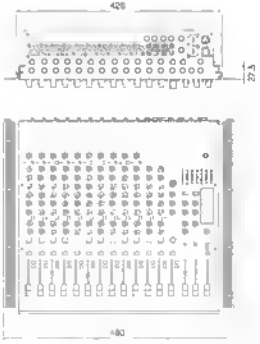
Output Connectors	Output Impedance	Appropriate Impedance	Rated Level	Max. Before Clipping	Connector Specifications
ST OUT (L, R)	150 Ω	600 Ω line	+4 dBu (1.23 V)	+24 dBu (12.3 V)	XLR-3-32 type (balanced) Phone jack (TRS) (balanced [T: hot, R: cold, S: ground])
GROUP OUT (1-4) AUX SEND (1-2) EFFECT SEND	150 Ω	10 kΩ line	+4 dBu (1.23 V)	+20 dBu (7.75 V)	Phone jack (TRS) (impedance balanced [T: hot, R: cold, S: ground])
CH INSERT OUT (CH# 1 to 8)	150 Ω	10 kΩ line	0 dBu (0.775 V)	+20 dBu (7.75 V)	Phone jack (TRS) (unbalanced [T: out, R: in, S: ground])
REC OUT (L, R)	600 Ω	10 kΩ line	-10 dBV (316 mV)	+10 dBV (3.16 V)	RCA pin jack
C-R OUT (L, R)	150 Ω	10 kΩ line	+4 dBu (1.23 V)	+20 dBu (7.75 V)	Phone jack (TRS) (impedance balanced [T: hot, R: cold, S: ground])
PHONES	100 Ω	40 Ω phone	3 mW	75 mW	Stereo phone jack

Where 0 dBu = 0.775 V and 0 dBV= 1 V

Specifications and descriptions in this owner's manual are for information purposes only. PYLE PRO reserves the right to change or modify products or specifications at any time, without prior notice.

Dimensional Diagrams

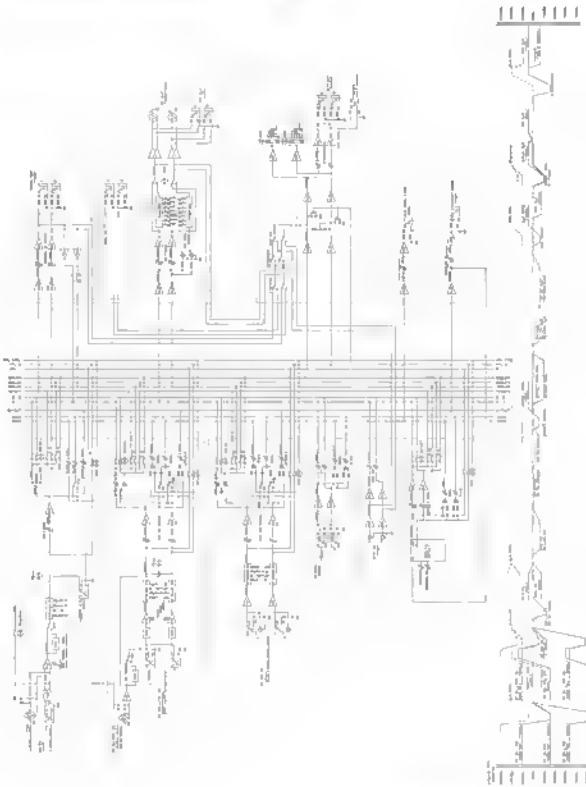




When mounted on rack

Unit: mm

Block Diagram and Level Diagram



Pyle pro, Inc.

1600 63rd Street

Brooklyn, NY 11204

www.pyleaudio.com